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81166 1290973908 HARMAN - BRINKS HOFER CHICAGO Brinks Hofer Gilson & Lione P.O. Box 10395 Chicago, II. 60610			EXAM	EXAMINER	
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Please find below and/or attached an Office communication concerning this application or proceeding.

The time period for reply, if any, is set in the attached communication.

Application No. Applicant(s) 10/556,232 HAULICK ET AL. Office Action Summary Examiner Art Unit DISLER PAUL 2614 -- The MAILING DATE of this communication appears on the cover sheet with the correspondence address --Period for Reply A SHORTENED STATUTORY PERIOD FOR REPLY IS SET TO EXPIRE 3 MONTH(S) OR THIRTY (30) DAYS. WHICHEVER IS LONGER, FROM THE MAILING DATE OF THIS COMMUNICATION. - Extensions of time may be available under the provisions of 37 CFR 1.136(a). In no event, however, may a reply be timely filed after SIX (6) MONTHS from the mailing date of this communication. If NO period for reply is specified above, the maximum statutory period will apply and will expire SIX (6) MONTHS from the mailing date of this communication - Failure to reply within the set or extended period for reply will, by statute, cause the application to become ABANDONED (35 U.S.C. § 133). Any reply received by the Office later than three months after the mailing date of this communication, even if timely filed, may reduce any earned patent term adjustment. See 37 CFR 1.704(b). Status 1) Responsive to communication(s) filed on 29 August 2008. 2a) This action is FINAL. 2b) This action is non-final. 3) Since this application is in condition for allowance except for formal matters, prosecution as to the merits is closed in accordance with the practice under Ex parte Quayle, 1935 C.D. 11, 453 O.G. 213. Disposition of Claims 4) Claim(s) 1-16: 19-23 is/are pending in the application. 4a) Of the above claim(s) _____ is/are withdrawn from consideration. 5) Claim(s) _____ is/are allowed. 6) Claim(s) 1-16:19-23 is/are rejected. 7) Claim(s) _____ is/are objected to. 8) Claim(s) _____ are subject to restriction and/or election requirement. Application Papers 9) The specification is objected to by the Examiner. 10) The drawing(s) filed on is/are; a) accepted or b) objected to by the Examiner. Applicant may not request that any objection to the drawing(s) be held in abevance. See 37 CFR 1.85(a). Replacement drawing sheet(s) including the correction is required if the drawing(s) is objected to. See 37 CFR 1.121(d). 11) The oath or declaration is objected to by the Examiner. Note the attached Office Action or form PTO-152. Priority under 35 U.S.C. § 119 12) Acknowledgment is made of a claim for foreign priority under 35 U.S.C. § 119(a)-(d) or (f). a) All b) Some * c) None of: Certified copies of the priority documents have been received. 2. Certified copies of the priority documents have been received in Application No. Copies of the certified copies of the priority documents have been received in this National Stage application from the International Bureau (PCT Rule 17.2(a)). * See the attached detailed Office action for a list of the certified copies not received.

Attachment(s)

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DETAILED ACTION

Response to Arguments

Applicant's arguments, filed 8/29/08, with respect to the rejection(s) of claim(s) 1
under June et al. in regard to the "processing the input signal by a beamformer" have
been fully considered and is non-persuasive.

Junes indeed disclose of a plurality of array wherein the input signals are processed by such a beamformer (fig.5A; par [0046-0048; 0054]/input signals with time-delay are summed for sound directivity denote such a beamforming).

 THIS ACTION IS MADE FINAL. Applicant is reminded of the extension of time policy as set forth in 37 CFR 1.136(a).

A shortened statutory period for reply to this final action is set to expire THREE MONTHS from the mailing date of this action. In the event a first reply is filed within TWO MONTHS of the mailing date of this final action and the advisory action is not mailed until after the end of the THREE-MONTH shortened statutory period, then the shortened statutory period will expire on the date the advisory action is mailed, and any extension fee pursuant to 37 CFR 1.136(a) will be calculated from the mailing date of the advisory action. In no event, however, will the statutory period for reply expire later than SIX MONTHS from the mailing date of this final action.

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Claim Rejections - 35 USC § 102

The following is a quotation of the appropriate paragraphs of 35 U.S.C. 102 that form the basis for the rejections under this section made in this Office action:

A person shall be entitled to a patent unless -

(e) the invention was described in (1) an application for patent, published under section 122(b), by another filled in the United States before the invention by the applicant for patent or (2) a patent granted on an application for patent by another filled in the United States before the invention by the applicant for patent, except that an international application filed under the treaty defined in section 351(a) shall have the effects for purposes of this subsection of an application filed in the United States only if the international application designated the United States and was published under Article 21(2) of such treaty in the English language.

 Claims 1,5; 14-15 are rejected under 35 U.S.C. 102(e) as being anticipated by June et al. (US 2003/0185410 A1).

Re claim 1, June et al. disclose of the Method for enhancing communication in a noisy environment comprising: receiving input signals emanating from at least two microphone arrays each comprising at least two microphones (fig.3 wt (201,202); par[0029]) and processing the input signals of each microphone array by a beamformer to determine temporal and spatial information about the input signals of each microphone array(fig.3, (304);par[0037,0039,0040,0042, 0046-0048,0032]/the only one microphone array (201) determined location of person in XY dimension as well as the temporal time info and as fig.3 (202) to help estimate the three dimensions of speaker and for efficiency each microphone array with fig.3 (308) may be used (par[0072]; fig.7).

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Re claim 5, the method according to one of the preceding claims, further comprising detecting speech activity for each microphone array (par[0042]).

Re claims 14-15 have been analyzed and rejected in view of claim 1.

Claim Rejections - 35 USC § 103

- 4. The following is a quotation of 35 U.S.C. 103(a) which forms the basis for all obviousness rejections set forth in this Office action:
 - (a) A patent may not be obtained though the invention is not identically disclosed or described as set forth in section 102 of this title, if the differences between the subject matter sought to be patented and the prior art are such that the subject matter as a whole would have been obvious at the time the invention was made to a person having ordinary skill in the art to which said subject matter pertains. Patentability shall not be negatived by the manner in which the invention was made.
- Claims 2-4, 6-11 are rejected under 35 U.S.C. 103(a) as being unpatentable over
 June et al. (US 2003/0185410 A1) and further in view Yang et al. (US 7,206,418 B2).

Re claim 2, the method according to claim 1 with processing each array of input signal with beamformer, however, June et al. wherein processing the input signals of each microphone array comprises processing by a wanted signal beamformer to obtain a wanted signal and by a blocking beamformer to obtain a blocking signal, preferably wherein the wanted signal beamformer is an adaptive beamformer, But, Yang et al. disclose of a system wherein the processing the input signals of each microphone array comprises processing by a wanted

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signal beamformer to obtain a wanted signal and by a blocking beamformer to obtain a blocking signal, preferably wherein the wanted signal beamformer is an adaptive beamformer (fig.2 wt (212); fig.3; col.5 line 1-37) for purpose of suppressing noise signal included in the speech signal array. Thus ,taking the combined teaching of June et al. and Yang et al. as a whole, it would have been obvious for one of the ordinary skill in the art at the time of the invention to have modify June et al. with the processing the input signals of each microphone array comprises processing by a wanted signal beamformer to obtain a wanted signal and by a blocking beamformer to obtain a blocking signal, preferably wherein the wanted signal beamformer is an adaptive beamformer for purpose of suppressing noise signal included in the speech signal array.

Re claim 3, the method according to claim 2, wherein processing the input signals of each microphone array further comprises deciding whether a signal is transmitted from a wanted signal direction, wherein the wanted signal beamformer is an adaptive beamformer being adapted only if no signal is transmitted from the wanted signal direction (Yang, col.11 line 36-41/signal may be adapted during time of non speech activity; col.10 line 44-51; fig.5 with col.8 line 5-20).

RE claim 4, the method according to claim 3, wherein deciding comprises determining a wanted signal power and a blocking signal

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power, wherein the wanted signal beamformer is adapted with the wanted signal and blocking signal power (fig. 2 wt (214a, fig. 3A); col.8 line 35-50/wanted and blocking signal power with a ratio or constant time the other). But, the combined teaching of June et al. and Yang et al. as a whole, fail to disclose of the specific wherein the wanted signal beamformer is adapted only if the blocking signal power is larger than a predetermined constant times the wanted signal power. But, the concept of having the specific wherein the wanted signal beamformer is adapted only if the blocking signal power is larger than a predetermined constant times the wanted signal power is simply the designer's need. Thus, it would have been obvious for one of the ordinary skill in the art to have modified the combined teaching of June et al. and Yang et al. as a whole, with the specific wherein the wanted signal beamformer is adapted only if the blocking signal power is larger than a predetermined constant times the wanted signal power for providing improve audio signal in presence of noise.

Re claim 6, the method according to claim 5, But, June et al. fail to disclose of the specific wherein detecting speech activity for a microphone array comprises determining a wanted signal power, a blocking signal power, and a background noise signal power comparing the wanted signal power with the blocking signal power and the background noise signal power. But, Yang et al. disclose of a system wherein detecting speech activity for a microphone array

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comprises determining a wanted signal power, a blocking signal power, and a background noise signal power comparing the wanted signal power with the blocking signal power and the background noise signal power (col.8 line 30-50/wanted and all noise power spectrums include background). Thus, taking the combined teaching of June et al. and Yang et al. as a whole, it would have been obvious for one of the ordinary skill in the art to have modified June et al. with the detecting speech activity for a microphone array comprises determining a wanted signal power, a blocking signal power, and a background noise signal power comparing the wanted signal power with the blocking signal power and the background noise signal power for enabling the device to adapt at the appropriate times.

RE claim 7, the method according to claim 6, wherein determining the wanted signal power with the microphone array (fig.3a; fig.2 (232a)/wanted signal power is determined), But, the combined teaching of June et al. and Yang et al. as a whole, fail to disclose of the specific wherein comparing the signal power of at least two microphone arrays and determining the highest power. But, the concept of comparing the signal power of at least two microphone arrays and determining the highest power is simply the designer's need. Thus, it would have been obvious for one of the ordinary skill in the art to have modified June et al. and Yang et al. as a whole, with such comparing the signal power of at least two microphone arrays and

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determining the highest power for enabling the best speech signals to adapt during period of speech activity.

Re claim 8, the method according to claim 5, but, June et al. fail to disclose of the specific wherein applying an attenuation to the processed input signals of a microphone array if no speech activity is detected for the microphone array. But, Yang et al. disclose of system wherein applying a scaling to the processed input signals of a microphone array if no speech activity is detected for the microphone array (fig.2 wt (236); col.8 line 20-35). Thus, taking the combined teaching of June et al. and Yang et al. as a whole, it would have been obvious for one of the ordinary skill in the art to have modified June et al. with the applying a scaling to the processed input signals of a microphone array if no speech activity is detected for the microphone array for purpose of obtaining a better estimate of the spectrum of the signal.

While, the combined teaching of June et al. and Yang et al. as a whole, disclose of the scaling the processed signals. But, they fail to disclose of the specific wherein applying an attenuation to the signals. But, applying an attenuation the signals is the designer's need, thus it would have been obvious for one of the ordinary skill in the art to have modified the combined teaching of June et al. and Yang

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et al. as a whole, with having the attenuation to the input processed signals for obtaining the sound spectrum.

Re claim 9, the method according to claim 8, wherein applying the attenuation is performed adaptivety, But, the combined teaching of June et al. and Yang et al. as a whole, fail to disclose of such wherein preferably by varying the attenuation in predetermined time steps between zero attenuation and a predetermined maximum attenuation. But, the concept of wherein preferably by varying the attenuation in predetermined time steps between zero attenuation and a predetermined maximum attenuation is simply the designer's preference. Thus, it would have been obvious for one of the ordinary skill in the art to have modified June et al. and Yang et al. as a whole, with such wherein preferably by varying the attenuation in predetermined time steps between zero attenuation and a predetermined maximum attenuation for obtaining a better estimate of the spectrum of the signal.

Re claim 10, the method according to one of the preceding claims claim 1, wherein processing comprises determining a gain control of the input signals for each microphone array(see claim 8).

Re claim 11, the method according to claim 10, wherein determining a qain control is performed adaptively (see claim 8).

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 Claims 12-13 are rejected under 35 U.S.C. 103(a) as being unpatentable over June et al. (US 2003/0185410 A1).

Re claim 12, the method according to claim 1, but, June et al. fail to disclose of the specific of further comprising selecting at least one output channel out of at least two output channels on which the processed signals are to be output. but, official notice is taken having the selecting at least one output channel out of at least two output channels on which the processed signals are to be output is well known in the art. Thus, it would have been obvious for one of the ordinary skill in the art to have modified June et al. with incorporating the selecting at least one output channel out of at least two output channels on which the processed signals are to be output for enabling the receipt of such sound processed signals.

Re claim 13, the method according to claim 12, but, June et al. fail to disclose of the specific of further wherein selecting the at least one output channel comprises determining an amplification for each selected output channel. But, official notice is taken having specific of further wherein selecting the at least one output channel comprises determining an amplification for each selected output channel is well known in the art. Thus, it would have been obvious for one of the ordinary skill in the art to have modified June et al. with

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incorporating the specific of further wherein selecting the at least one output channel comprises determining an amplification for each selected output channel for enabling the receipt of such sound processed signals.

 Claims 16; 19-22 are rejected under 35 U.S.C. 103(a) as being unpatentable over June et al. (US 2003/0185410 A1) and Yang et al. and Roddy (US 6,363,156).

Re claim 16, June et al. disclose of the Communication system comprising: at least two microphone arrays each comprising at least two microphones to produce microphone signals, at least one analog/digital converter having an input for receiving said microphone signals and an output for providing digital microphone signals (fig.3 wt (310); par[0037])), digital signal processing means having an input for receiving the digital microphone signals, being configured to process the digital microphone signals of each microphone array by a beamformer to determine temporal and spatial information about the microphone signals of each microphone array, and having an output to provide processed output signals determined location of person in XY dimension as well as the temporal time info and as fig.3 (202) to help estimate the three dimensions of source) and having an output to provide processed output signals (fig.3 wt (320); par[0038]).

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However, June et al. fail to disclose of the specific wherein the processed output signals to at least two loudspeakers. But, Roddy disclose of a spatial enhancement audio with microphone input signal system wherein the processed output signals to at least two loudspeakers (fig.1-2 wt (30,34); col.2 line 35-65) for the purpose of enabling the passenger in a vehicle to hear the speech of the drivers or others in the vehicle. Thus, taking the combined teaching of June et al. and Roddy as a whole, it would have been obvious for one of the ordinary skill in the art at the time of the invention to have modify June et al. with the processed output signals to at least two loudspeakers for the purpose of enabling the passenger in a vehicle to hear the speech of the drivers or others in the vehicle.

The combined teaching of June et al. and Roddy as a whole, further disclose of the where the digital signal processing means is further configured to detect speech activity through each microphone array (par [0038-0040]/speech to be detected by sensor mic).

But, the combined teaching of June et al. and Roddy as a whole, fail to disclose of the specific where the digital signal processing means is further configured to determine and apply an attenuation to the processed digital microphone signals of one of the microphone array if no speech is detected by that microphone array. But, Yang et al. disclose of system wherein applying a scaling to the processed input signals of a microphone array if no speech activity is detected

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for the microphone array (fig.2 wt (236); col.8 line 20-35). Thus, taking the combined teaching of June et al. and Roddy and Yang et al. as a whole, it would have been obvious for one of the ordinary skill in the art to have modified June et al. with the applying a scaling to the processed input signals of a microphone array if no speech activity is detected for the microphone array for purpose of obtaining a better estimate of the spectrum of the signal.

While, the combined teaching of June et al. and Roddy and Yang et al. as a whole, disclose of the scaling the processed signals. But, they fail to disclose of the specific wherein applying an attenuation to the signals. But, applying an attenuation the signals is the designer's need, thus it would have been obvious for one of the ordinary skill in the art to have modified the combined teaching of June et al. and Yang et al. as a whole, with having the attenuation to the input processed signals for obtaining the sound spectrum.

Re claim 19, the communication system according to claim 16, wherein the digital signal processing means is further configured to select at least one loudspeakers out of the at least two loudspeakers on which the processed signals are to be output (col.3 line 1-5 & line 14-16/signals from microphones to appropriate/selected speakers).

Re claim 20, a vehicle cabin comprising a communication system according to one of the claims 16 or 19, and at least two

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loudspeakers, wherein each microphone array and each loudspeaker is associated with a passenger seat (Roddy, fig.1, col.2 line 40-50).

Re claim 21 which is a broader claim limitation of claim 16, has been analyzed and rejected with respect to such claim 16.

Re claim 22, the communication system according to claim 21, where the spatial information includes spatial information about a plurality of signal sources (par [0011, 0014]/plurality of sound sources locations for spatial info to be determined).

 Claim 23 is rejected under 35 U.S.C. 103(a) as being unpatentable over June et al. (US 2003/0185410 A1) and Yang et al. and Roddy (US 6,363,156) and Higuchi et al. (US 5,721,771).

Re claim 23, the communication system according to claim 21 with the digital processing, But, the combined teaching of June et al. and Yang et al. and Roddy as a whole, fail to disclose of wherein the processing means is further configured to detect and overdrive that reduce feedback effects. But, Higuchi et al. disclose of a system wherein such concept of processing means is further configured to detect and overdrive that reduce feedback effects (fig.1 wt (11); col.6 line 11-22; col.5 line 55-63). Thus, taking the combined teaching of June et al. and Yang et al. and Roddy and Higuchi et al.

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as a whole, it would have been obvious for one of the ordinary skill in the art at the time of the invention to have modified the combined teaching of June et al. and Yang et al. and Roddy as a whole, with incorporating the system wherein such concept of processing means is further configured to detect and overdrive that reduce feedback effects for purpose completely eliminating acoustic echo for improving speech signal.

Conclusion

Any inquiry concerning this communication or earlier communications from the examiner should be directed to Disler Paul whose telephone number is 571-270-1187. The examiner can normally be reached on 7:30-5:00.

If attempts to reach the examiner by telephone are unsuccessful, the examiner's supervisor, Chin Vivian can be reached on 571-272-7848. The fax phone number for the organization where this application or proceeding is assigned is 571-273-8300.

Information regarding the status of an application may be obtained from the Patent Application Information Retrieval (PAIR) system. Status information for published applications may be obtained from either Private PAIR or Public PAIR. Status information for unpublished applications is available through Private PAIR only. For more information about the PAIR system, see http://pair-direct.uspto.gov. Should you have questions on access to the Private PAIR system, contact the Electronic Business Center (EBC) at 866-217-9197 (toll-free). If you would like assistance from a USPTO Customer Service Representative or access to the automated information system, call 800-786-9199 (IN USA OR CANADA) or 571-272-1000.

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/D. P./ Examiner, Art Unit 2614

> /Xu Mei/ Primary Examiner, Art Unit 2614